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Please find below and/or attached an Office communication concerning this application or proceeding.

The time period for reply, if any, is set in the attached communication.

Notice of the Office communication was sent electronically on above-indicated "Notification Date" to the following e-mail address(es):

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DETAILED ACTION

Claim Objections

1. Claim 1 is objected to because of the following informalities: In the phrase "convert the the one or more..." the second "the" is extraneous. Appropriate correction is required.

Claim Rejections - 35 USC § 102

2. The following is a quotation of the appropriate paragraphs of 35 U.S.C. 102 that form the basis for the rejections under this section made in this Office action:

A person shall be entitled to a patent unless –

(b) the invention was patented or described in a printed publication in this or a foreign country or in public use or on sale in this country, more than one year prior to the date of application for patent in the United States.

3. Claims 1-35 are rejected under 35 U.S.C. 102(b) as being anticipated by Harada (6,038,529).

As per claim 1, Harada teaches an apparatus for use in transmitting digital data through an audio channel the apparatus comprising:

a data coder configured to convert the digital data into one or more types of sound parameters; and a sound synthesizer coupled to the data coder and configured to convert the one or more types of sound parameters into acoustic sound waves to acoustically transfer the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 2, Harada teaches the apparatus of claim 1, further comprising: a storage medium configured to store one or more sets of relationships between bit patterns and one or more types of sound parameters; and wherein the data coder is configured to convert the digital data into the one or more types of sound parameters in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 3, Harada teaches the apparatus of claim 2, wherein the storage medium comprises a look up table that predefines the one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Col.9, line 24 – Col.10, line 62).

As per claim 4, Harada teaches the apparatus of claim 1, wherein a sound parameter represents one value or a range of values representative of user authentication information (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 5, Harada teaches the apparatus of claim 1, wherein the one or more sound parameters comprises at least one speech parameter representative of artificial speech (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 6, Harada teaches an apparatus for use in receiving digital data through an audio channel, the apparatus comprising: a sound analyzer configured to receive sound and to extract one or more types of sound parameters from the received sound; and a data decoder coupled to the sound analyzer and configured to convert the extracted one or more types of sound parameters into the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 7, Harada teaches the apparatus of claim 6, further comprising: a storage medium configured to store one or more sets of relationships between bit patterns and one or more types of sound parameters; and wherein the data decoder is configured to convert the extracted one or more types of sound parameters into the digital data in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 8, Harada teaches the apparatus of claim 7, wherein the storage medium comprises a look up table that predefines one or more sets of relationships between the bit patterns and the one or more types of sound patterns (Col.9, line 24 – Col.10, line 62).

As per claim 9, Harada teaches the apparatus of claim 6, wherein a sound parameter represents one value or a range of values representative of user authentication information (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 10, Harada teaches the apparatus of claim 6, wherein the extracted one or more sound parameters comprise at least one speech parameter representative of artificial speech (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 11, Harada teaches a method for use in transmitting digital data through an audio channel, the method comprising: converting digital data to be transmitted into one or more types of sound parameters; and converting the one or more sound parameters into acoustic sound waves to acoustically transfer the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 12, Harada teaches the method of claim 11, further comprising: storing one or more sets of relationships between bit patterns and one or more types of sound parameters; and wherein converting digital data to be transmitted comprises converting the digital data into the one or more types of sound parameters in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 13, Harada teaches the method of claim 12, wherein storing the one or more sets of relationships comprises storing a look up table that predefines one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Col.9, line 24 – Col.10, line 62).

As per claim 14, Harada teaches the method of claim 11, wherein a sound parameter represents one value or a range of values representative of user authentication information (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 15, Harada teaches the method of claim 11, wherein the one or more sound parameters comprises at least speech parameter representative of artificial speech (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 16, Harada teaches a method for use in receiving digital data through an audio channel, the method comprising: extracting one or more types of sound parameters from received acoustic sound waves; and converting the extracted one or more types of sound parameters into the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 17, Harada teaches the method of claim 16, further comprising: storing one or more sets of relationships between bit patterns and the one or more types of sound parameters; and wherein converting the extracted one or more types of sound parameters comprises converting the extracted one or more types of sound parameters into the digital data in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 18, Harada teaches the method of claim 17, wherein storing the one or more sets of relationships comprises storing a look up table that predefines the one or more sets of relationships (Col.9, line 24 – Col.10, line 62).

As per claim 19, Harada teaches the method of claim 16, wherein a sound parameter represents one value or a range of values representative of user authentication information (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 20, Harada teaches the method of claim 16, wherein the extracted one or more sound parameters comprise at least one speech parameter representative of artificial speech (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 21, Harada teaches an apparatus for use in transmitting digital data through an audio channel, the apparatus comprising: means for converting digital data to be transmitted into one or more types of sound parameters; and means for converting the one or more types of sound parameters into acoustic sound waves to acoustically transfer the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 22, Harada teaches the apparatus of claim 21, further comprising: means for storing one or more sets of relationships between bit patterns and one or more types of sound parameters; and wherein the means for converting converts the digital data into the one or more types of sound parameters in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 23, Harada teaches the apparatus of claim 22, wherein the means for storing stores a look up table that predefines one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Col.9, line 24 – Col.10, line 62).

As per claim 24, Harada teaches an apparatus for use in receiving digital data through an audio channel, the apparatus comprising: means for extracting one or more types of sound parameters from received acoustic sound waves; and means for converting the extracted one or more types of sound parameters into the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 25, Harada teaches the apparatus of claim 24, further comprising: means for storing one or more sets of relationships between bit patterns and one or more types of sound parameters; and wherein the means for converting converts the extracted one or more types of sound parameters into the digital data in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 26, Harada teaches the apparatus of claim 25, wherein the means for storing stores a look up table that predefines the one or more sets of relationships between the one or more types of sound parameters (Col.9, line 24 – Col.10, line 62).

As per claim 27, Harada teaches a machine readable medium used for transmitting digital data through an audio channel, the machine readable medium comprising: codes for converting digital data to be transmitted into one or more types of sound parameters; and codes for converting the one or more types of sound parameters into acoustic sound waves to acoustically transfer the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 28, Harada teaches the medium of claim 27, further comprising: one or more sets of relationships between bit patterns and one or more types of sound parameters; and wherein the codes for converting converts the digital data into the one or more types of sound parameters in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 29, Harada teaches a machine readable medium used for receiving digital data through an audio channel, the machine readable medium comprising: codes for extracting one or more types of sound parameters from received compressed sound; and codes for converting the extracted one or more types of sound parameters into the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 30, Harada teaches the medium of claim 29, further comprising: one or more sets of relationships between bit patterns and one or more types of sound

parameters; and wherein the codes for converting converts the extracted one or more types of sound parameters into the digital data in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 31, Harada teaches the apparatus for use in transmitting and receiving digital data through an audio channel, the apparatus comprising: means for converting digital data to be transmitted into one or more types of sound parameters; means for generating acoustic sound waves based on the one or more types of sound parameters; means for extracting one or more types of sound parameters from received acoustic sound waves; and means for converting the extracted one or more types of sound parameters into the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 32, Harada teaches the apparatus of claim 31, further comprising: means for storing one or more sets of relationships between bit patterns and one or more types of sound parameters; and wherein the means for converting converts the digital data into the one or more types of sound parameters in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound parameters, and wherein the means for converting converts the extracted one or more types of sound parameters into the digital data in accordance with the one or more sets of relationships between the bit patterns and the one or more types of sound parameters (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 33, Harada teaches the apparatus of claim 32, wherein the means for storing stores a look up table that predefines one or more sets of relationships

between the bit patterns and the one or more types of sound parameters (Col.9, line 24 – Col.10, line 62).

As per claim 34, Harada teaches a processor for use in transmitting digital data through an audio channel, the processor comprising: a processor circuit configured to :convert digital data to be transmitted into one or more types of sound parameters, and converting the one or more types of sound parameters into acoustic sound Waves to acoustically transfer the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

As per claim 35, Harada teaches a processor for use in receiving digital data through an audio channel, the processor comprising: a processing circuit configured to: extract one or more types of sound parameters from received acoustic sound waves, and, convert the extracted one or more types of sound parameters into the digital data (Figs.3-4, Col.4. line 51-Col.5, line 52).

Response to Arguments

4. Applicant's arguments with respect to claims 1-35 have been considered but are moot in view of the new ground(s) of rejection.

Conclusion

5. Applicant's amendment necessitated the new ground(s) of rejection presented in this Office action. Accordingly, **THIS ACTION IS MADE FINAL**. See MPEP § 706.07(a). Applicant is reminded of the extension of time policy as set forth in 37 CFR 1.136(a).

A shortened statutory period for reply to this final action is set to expire **THREE MONTHS** from the mailing date of this action. In the event a first reply is filed within **TWO MONTHS** of the mailing date of this final action and the advisory action is not mailed until after the end of the **THREE-MONTH** shortened statutory period, then the shortened statutory period will expire on the date the advisory action is mailed, and any extension fee pursuant to 37 CFR 1.136(a) will be calculated from the mailing date of the advisory action. In no event, however, will the statutory period for reply expire later than **SIX MONTHS** from the date of this final action.

Any inquiry concerning this communication or earlier communications from the examiner should be directed to Vijay B. Chawan whose telephone number is (571) 272-7601. The examiner can normally be reached on Monday Through Friday 6:30-3:00.

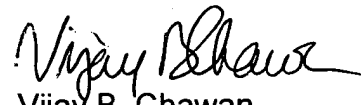
If attempts to reach the examiner by telephone are unsuccessful, the examiner's supervisor, Richemond Dorvil can be reached on (571) 272-7602. The fax phone

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number for the organization where this application or proceeding is assigned is (571) 273-8300.

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Vijay B. Chawan
Primary Examiner
Art Unit 2626

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PRIMARY EXAMINER**

vbc
12/9/07